

Brána se konfiguruje přes www rozhraní a pomocí klávesnice připojeného telefonu. Původní heslo je "admin".

Sip Server a Outbound Proxy: zjistíte na <http://www.802.cz/zjistení-sip-serveru/>

Grandstream Device Configuration

STATUS **BASIC SETTINGS** **ADVANCED SETTINGS**

MAC Address: 00.0B.82.03.FE.08

WAN IP Address:

Product Model: HT286

Software Version: Program-- 1.0.5.18 Bootloader-- 1.0.0.21 HTML-- 1.0.0.42 VOC-- 1.0.0.7
detected NAT type is open Internet

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STATUS **BASIC SETTINGS** **ADVANCED SETTINGS**

End User Password: (end user password to configure this device)

IP Address: dynamically assigned via DHCP (default) or PPPoE
(will attempt PPPoE if DHCP fails and following is non-blank)

PPPoE account ID:

PPPoE password:

Preferred DNS server: . . .

statically configured as:

IP Address: . . .

Subnet Mask: . . .

Default Router: . . .

DNS Server 1: . . .

DNS Server 2: . . .

Time Zone: ▾

Daylight Savings Time: No Yes (if set to Yes, display time will be 1 hour ahead of normal time)

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Grandstream Device Configuration

STATUS

BASIC SETTINGS

ADVANCED SETTINGS

Admin Password: (admin password to configure this device)
SIP Server: (e.g., sip.mycompany.com, or IP address)
Outbound Proxy: (e.g., proxy.myprovider.com, or IP address, if any)
SIP User ID: (the user part of an SIP address)
Authenticate ID: (can be identical to or different from SIP User ID)
Authenticate Password:
Name: (optional, e.g., John Doe)

Advanced Options:

Preferred Vocoder: (in listed order)
choice 1:
choice 2:
choice 3:
choice 4:
choice 5:
choice 6:
choice 7:

G723 rate: 6.3kbps encoding rate 5.3kbps encoding rate

iLBC frame size: 20ms 30ms

iLBC payload type: (between 96 and 127, default is 98)

Silence Suppression: No Yes

Voice Frames per TX: (up to 10/20/32/64 for G711/G726/G723/other codecs respectively)

Layer 3 QoS: (Diff-Serv or Precedence value)

Layer 2 QoS: 802.1Q/VLAN Tag 802.1p priority value (0-7)

Use DNS SRV: No Yes

User ID is phone number: No Yes

SIP Registration: Yes No

Unregister On Reboot: Yes No

Register Expiration: (in minutes, default 1 hour, max 45 days)

Early Dial: No Yes (use "Yes" only if proxy supports 484 response)

Dial Plan Prefix: (this prefix string is added to each dialed number)

No Key Entry Timeout: (in seconds, default is 4 seconds)

Use # as Dial Key: No Yes (if set to Yes, "#" will function as the "(Re-)Dial" key)

local SIP port: (default 5060)

local RTP port: (1024-65535, default 5004)

Use random port: No Yes

NAT Traversal: No
 Yes, STUN server is: (URI or IP:port)

keep-alive interval: (in seconds, default 20 seconds)

Use NAT IP (if specified, this IP address is used in SIP/SDP message)

Proxy-Require: (if specified, the content will appear in Proxy-Require header)

TFTP Upgrade Server: . . . (for remote software upgrade and configuration)

HTTP Upgrade Server: (IP address or URL)

Auto Upgrade: No Yes, check for new firmware every days (default 7 days)

SUBSCRIBE for MWI: No, do not send SUBSCRIBE for Message Waiting Indication
 Yes, send periodical SUBSCRIBE for Message Waiting Indication

Offhook Auto-Dial: (User ID/extension to dial automatically when offhook)

Enable Call Features: No Yes (if Yes, Call Forwarding & Call-Waiting-Disable are supported locally)

Disable Call-Waiting: No Yes

Send DTMF: in-audio via RTP (RFC2833) via SIP INFO

DTMF Payload Type:

Send Flash Event: No Yes (Flash will be sent as a DTMF event if set to Yes)

FXS Impedance: ▾

NTP Server: (URI or IP address)

Send Anonymous: No Yes (caller ID will be blocked if set to Yes)

Lock keypad update: No Yes (configuration update via keypad is disabled if set to Yes)

Update

Cancel

Reboot